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   QUESTION 1Refer to the exhibit. When users call number 00044156789453, they have to wait for several seconds before the call is sent.

 Which action sends the call immediately? A.   In the translation pattern, change the CSS.B.   In the route pattern, check the Allow Device Override check box.C.   In the route pattern, check the Urgent Priority check box.D.   In the translation pattern, check the Route Next Hop By Calling Party Number check box. Answer: C QUESTION 2Which three options are valid SIP URI formats? (Choose three.) A.   jsmith@cisco.comB.   tel:98765@cisco.comC.   phone:98765@cisco.comD. sip:12345@10.10.10.10:5060;user=phoneE.   sip:bob.cisco.com Answer: ABD QUESTION 3Which two codecs are the best codecs to use in WAN and LAN? (Choose two.) A.   WAN codec G.711B.   LAN codec G.729C.   WAN codec G.729D.   LAN codec

G.711 Answer: CD QUESTION 4How do you configure calling privileges on Cisco Unified Communications Manager Express? A. corlistB. translation profileC. zonesD. access listE. search rules Answer: A QUESTION 5Which DiffServ PHB classification is recommended by Cisco for interactive video traffic? A. AF31B. CS3C. EFD. AF41E. CS4 Answer: D QUESTION 6Which three descriptions are true? (Choose three.) A. The class of service is a parameter that is used for traffic classification and marking at Layer 2.B. The IP precedence is a parameter that is used for traffic classification and marking at Layer 2.C. The differentiated services code point is a parameter that is used for traffic classification and marking at Layer 2.D. The class of service is a parameter that is used for traffic classification and marking at Layer 3.E. The IP precedence is a parameter that is used for traffic classification and marking at Layer 3.F. The differentiated services code point is a parameter that is used for traffic classification and marking at Layer 3. Answer: AEF QUESTION 7What are the two benefits of using SIP dial rules on an IP phone? (Choose two) A. The phone can initiate dialing without sending any signaling messages to Cisco Unified Communications Manager.B. The phone can detect invalid numbers and play a reorder tone without sending any signaling messages to Cisco Unified Communications Manager.C. If dialed digits match an entry of a SIP dial rule, the dialed string is sent in a single SIP 200 OK message to Cisco Unified Communications Manager.D. If Cisco Unified Communications Manager requires more digits, KPML can be used to send the remaining digits from the SIP phone to Cisco Unified Communications Manager one-by-one.E. If Cisco Unified Communications Manager requires more digits, another en bloc message is used to send the remaining digits from the SIP. Answer: BD QUESTION 8Which three Cisco Unified CallManager configuration steps are required to support third party SIP phones? (Choose three.) A. configure the device in Cisco Unified CallManagerB. change the proxy address in the SIP phone to an IP address or the Fully Qualified Domain Name (FQDN) of Cisco Unified CallManagerC. associate the device with the end userD. configure the phone with the TLS username and passwordE. configure the end user in Cisco Unified CallManagerF. add the MAC address of the Cisco Unified CallManager server to the SIP phone configuration page Answer: ACE QUESTION 9 ????????   **100% Full Money Back Guarantee Promised By Braindump2go to All 300-070 Exam Candiates: Braindump2go is confident that our NEW UPDATED 300-070 Exam Questions and Answers are changed with Cisco Official Exam Center, If you cannot PASS 300-070 Exam, nevermind, we will return your full money back! Visit Braindump2go exam dumps collection website now and download 300-070 Exam Dumps Instantly Today!**

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